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TITLE: A DISTORTION COMPENSATION CIRCUIT, AND
 A TRANSMISSION APPARATUS INCLUDING THE
 SAME

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A DISTORTION COMPENSATION CIRCUIT, AND A TRANSMISSION APPARATUS INCLUDING THE SAME

Background of the Invention

5 1. Field of the Invention

This invention relates to a distortion compensation circuit for compensating distortion in a power amplifier and particularly to a distortion compensation circuit with so-called Memory-effect and to a transmission apparatus to which the distortion compensation circuit is
10 applicable.

2. Description of the Related Art

Recently, in cellular phone systems, new systems have been developed such as a W-CDMA (Wideband-Code Division Multiple Access), CDMA (Code Division Multiple Access) 2000, which are called as the third
15 generation system. In these cellular phone systems, an occupied bandwidth per one channel becomes considerably wider than that in a conventional PDC (Personal Digital Cellular) and PHS (Personal Handyphone System: Trademark). More specifically, the above-described occupied bandwidth per one channel is, for example, in the PDC,
20 about 20 KHz and, in the PHS, it is about 200 KHz. Further, the occupied bandwidth in the CDMA 2000 is about 1.2 MHz and that in the W-CDMA becomes about 4 MHz.

Here, if a signal having a wide bandwidth, such as signals handled in the CDMA 2000 and the W-CDMA, is amplified, it becomes
25 difficult to keep the characteristics of the amplifier within the band flat. Particularly, since requirements for distortion in a power amplifier (PA) is severe, designing of such power amplifier tends to be difficult. On the other hand, the distortion compensation technology is useful for improving distortion in a power amplifier and improving the efficiency
30 thereof. However, it is difficult to efficiently compensate distortion in power amplifiers having a large deviation over its band. Thus, there is

no appropriate method of compensating distortion in such power amplifier.

On the other hand, as an example of technology for compensating distortion in a power amplifier having the so-called Memory-effect, a Predistortion method is disclosed in which a predistortion signal is generated in a baseband section with a DSP (Digital Signal Processor). For example, J. Kim and K. Konstantinou "Digital predistortion of wideband signals based on power amplifier model with memory", IEEE Electronics letters, 8th November 2001, Vol. 37, No. 23, pp. 1417-1418 discloses this.

According to the technology disclosed in this document, a signal voltage at a sample point is compared with a signal voltage at a previous sample point by one sample with regard to a baseband signal sequence, and a predistortion signal generation section generates a predistortion signal on the basis of the comparison result.

However, in this method of generating a predistortion signal, because the predistortion signal is generated at the baseband section, it is difficult to compensate distortion over the entire bandwidth of the power amplifier. In addition, because the predistortion signal is generated by comparison with only one-sampling previous value, this method has no effect on power amplifiers showing the strong memory effect.

Summary of the Invention

An aspect of the present invention provides a distortion compensation circuit through a predistortion technology effective for power amplifiers with the memory effect and a transmission apparatus including the same.

According to the present invention, the present invention provides a distortion compensation circuit includes an A/D converter for digitizing a signal voltage value of a signal after quadrature modulating a baseband signal, a subtractor supplied with the output data of the A/D converter, a voltage value data output section for outputting a voltage value data

corresponding to the output data of the subtractor by selecting from a plurality of pieces of previously stored voltage value data, an amplitude impulse response accumulation adding section for supplying, to the subtractor in accordance with the voltage value data from the voltage value data outputting section, an accumulation adding value of values obtained by multiplying the signal voltage value after quadrature modulation by impulse response values corresponding to amplitude characteristic of the power amplifier, a D/A converter for converting the voltage value data from the voltage value data outputting section into an analog signal as an output predistortion signal regarding the amplitude component of the power amplifier.

Another aspect of the present invention provides a transmission apparatus includes a quadrature modulation section for quadrature-modulating a transmission baseband signal, a distortion compensation circuit according to the present invention, and a conversion/removing section supplied with the output signal from the distortion compensation circuit for effecting frequency conversion and removal of electromagnetic interference (EMI).

In still another aspect of the present invention, the predistortion signal for compensating distortion in a power amplifier is generated from the signal after quadrature modulation of the baseband signal.

Still further aspect of the present invention provides compensation of distortion over the entire bandwidth of a power amplifier with the memory effect.

25

Brief Description of the Drawings

The object and features of the present invention will become more readily apparent from the following detailed description taken in conjunction with the accompanying drawings in which:

30 Fig. 1 is a graphical drawing illustrating input/output characteristics of a power amplifier with the memory effect used in the

transmission apparatus according to the present invention;

Fig. 2 is a graphical drawing illustrating AM/AM characteristics of the power amplifier with the memory effect shown in Fig. 1;

Fig. 3 is a graphical drawing illustrating an example of impulse
5 response of the power amplifier with the memory effect shown in Fig. 1;

Fig. 4 is a graphical drawings illustrating convolution integration using the impulse response shown in Fig. 3;

Fig. 5 is a graphical drawing illustrating an example of a curve of data stored in a lookup table used in a distortion compensation circuit
10 according to the present invention;

Fig. 6 is a block diagram of a transmission section of a portable telephone terminal using the distortion compensation circuit according to the present invention;

Fig. 7 is a graphical drawing illustrating a frequency relation
15 between the memory effect of the power amplifier and distortion compensation according to the represent invention;

Fig. 8 is a bock diagram of the distortion compensation section according to the present invention;

Fig. 9 is a graphical drawing illustrating a frequency component
20 of the output signal of the power amplifier without distortion compensation, obtained through simulation;

Fig. 10 is a graphical drawing illustrating a frequency component of the output signal of the power amplifier supplied with a predistortion signal obtained through simulation in consideration of the characteristic
25 curve po3 shown in Fig. 1 without consideration of the memory effect;

Fig. 11 is a graphical drawing illustrating a frequency component of the output signal of the power amplifier with distortion compensation according to the present invention, obtained through simulation;

Fig. 12 is a graphical drawing illustrating output power
30 dependencies of ACP_low for three cases shown in Figs. 9 to 11;

Fig. 13 is a graphical drawing illustrating output power

dependencies of ACP_high for three cases shown in Figs. 9 to 11;

Fig. 14 is a graphical drawing illustrating output power dependencies of deviation within the band; and

Fig. 15 is a graphical drawing illustrating output power dependencies of asymmetry in ACP.

The same or corresponding elements or parts are designated with like references throughout the drawings.

Detailed Description of the Preferred Embodiment

A preferred embodiment will be described with reference to the attached drawings.

Operation Principle of Distortion Compensation

First, an operation principle of distortion compensation according to the present invention will be described.

Assuming that a time function of an input signal voltage of a power amplifier is $u(t)$, and that a time function of an output signal voltage of the power amplifier is $y(t)$, an input/output characteristic of the power amplifier is generally given by:

$$y(t) = \int_1 h_1(\tau)u(t-\tau)d\tau + \int \int h_2(\tau_1, \tau_2)u(t-\tau_1)u(t-\tau_2)d\tau_1d\tau_2 + \dots \quad (1)$$

where $h_k(\tau_1, \dots, \tau_k)$ is k^{th} order Volterra kernel representing k^{th} order non-linear impulse response. In addition, n -multiple integral term (n is a natural number) in the Equation (1) provides an output of n^{th} order components of the input signal voltage $u(t)$. That is, for example, the quintuple integral term provides a fifth order component, an IM (Inter Modulation) 5 component, and a fundamental wave component and an IM 3 component generated from the fifth order term.

Next, the Equation (1) is subjected to Fourier transform to provide an Equation (2). That is, the above-described Equation (1) represents

the input/output characteristic of the power amplifier in time domain, and the Equation (2) represents the input/output characteristic of the power amplifier in frequency domain.

$$V_{out}(\omega) = H_1(\omega_1) \cdot V_{in}(\omega) + H_2(\omega_1, \omega_2) \cdot V_{in}(\omega)^2 + H_3(\omega_1, \omega_2, \omega_3) \cdot V_{in}(\omega)^3 + \dots \quad (2)$$

where $V_{in}(\omega)$ represents Fourier transform of the input signal voltage $u(t)$, and $V_{out}(\omega)$ represents Fourier transform of the output signal voltage $y(t)$.

In addition, in the Equation (2), ω_1 , ω_2 , and ω_3 represent frequency components of the input signal V_{in} to be supplied to the power amplifier. $H_n(\omega)$ represents Fourier transform of Volterra kernel. Here, $H_n(\omega)$ is generally a complex number.

Here, in the power amplifier of which distortion is to be compensated, the memory effect means in the Equation (2) that the frequency dependency of $H_n(\omega)$ is so large that it cannot be neglected. Thus, the Equation (2) represents both of so-called AM/AM and AM/PM characteristics.

Thus, in the condition that an input signal voltage to the power amplifier is such that the characteristic of the power amplifier is linearized by distortion compensation, Equation (3) or (4) is to be satisfied. In these Equations (3) and (4), G_0 represents a linear gain of the power amplifier. Further, when the Equation (3) or (4) is satisfied, the power amplifier having the linear gain G_0 does not show any memory effect, and becomes a Memory-less power amplifier.

$$G_0 \cdot u(t) = \int_0^1 h_1(\tau) u(t - \tau) d\tau + \int \int h_2(\tau_1, \tau_2) u(t - \tau_1) u(t - \tau_2) d\tau_1 d\tau_2 + \dots \quad (3)$$

$$G_0 \cdot V_{in}(\omega) = H_1(\omega_1) \cdot V_{in}(\omega) + H_2(\omega_1, \omega_2) \cdot V_{in}(\omega)^2 + H_3(\omega_1, \omega_2, \omega_3) \cdot V_{in}(\omega)^3 + \dots \quad (4)$$

Therefore, the generation of the predistortion signal results in obtaining the input voltage $u(t)$ satisfying the Equation 3 or Equation 4.

Hereinafter, operation of distortion compensation according to the present invention will be described with an example.

5 Figs. 1 and 2 illustrate input/output characteristics of an exemplified power amplifier with the memory effect, i.e., a Memory-System. Five curves po1, po2, po3, po4, and po5 in Fig. 1 show input/output characteristics corresponding to respective frequencies f1 to f5 in Fig. 2. Here, since the input/output characteristics in Fig. 1 are
10 represented in power, they are converted to represent the input/output characteristics in voltage, and these five input/output characteristics are approximated with fifth order polynomial given by:

$$Vo(fi) = am1(fi) \cdot Vin(fi) + am3(fi) \cdot Vin^3(fi) + am5(fi) \cdot Vin^5(fi) \quad (5)$$

15

where $i = 1, 2, 3, 4, 5$. Then, smoothly connecting $am1(fi)$, $am3(fi)$, and $am5(fi)$ obtained by the Equation 5 in a direction of frequency provides respective curves shown in Fig. 2.

Here, from studying variation in time of the output voltage of the
20 power amplifier having the input/output characteristics represented by the Equation (5), $am1(fi)$, $am3(fi)$, and $am5(fi)$ in the Equation (5) can be considered as representation of filter characteristics. Therefore, the responses of these filters can be represented by convolution integration with impulse responses. Figs. 3 and 4 show this operation, where Fig. 3
25 shows a value $imp(i)$ of impulse response at every sampling point (sampling interval τ) and Fig. 4 shows the convolution integration with these impulse responses.

Further, if it is assumed that "i" and "k" in Figs. 3 and 4 are considered as sampling points by digitizing the signal in time base, the
30 output signal voltage $u(k)$ at a time k is an accumulation adding value of products of past input signal voltages $u(k-1)$, $u(k-2)$, ..., and respectively

corresponding impulse responses as given by:

$$y(k) = \sum_{i=1}^n u(k-i) \cdot \text{imp1}(i) + \sum_{i=1}^n u(k-i)^3 \cdot \text{imp3}(i) + \sum_{i=1}^n u(k-i)^5 \cdot \text{imp5}(i) \quad (6)$$

- 5 where $\text{imp}(i)$ in the Equation (6) represents an impulse response of $\text{am}(i)$ in the Equation (5). Further, "n" in the Equation (6) is a natural number, for example, $n = 4$.

Here, if the power amplifier would not have the memory effect, that is, it would be a Memory-Less System, because the duration of the
10 impulse response is extremely short, the above-described accumulation adding value is negligible. However, if the power amplifier has the memory effect, that is, it is a Memory System, the above-described accumulation adding value adversely effects. Particularly, the third order and fifth order terms in the Equation (6) represent distortion,
15 which is influenced from accumulated past distortion.

The above-description provides time domain behavior if the power amplifier has the memory effect, i.e., it is a Memory System.

In consideration of these facts, the distortion compensation circuit according to the present invention provides distortion compensation of
20 the power amplifier as follows:

A predistortion signal for distortion compensation for a power amplifier, that is, the input signal voltage $u(k)$ to the power amplifier satisfies Equations (7-1), (7-2), and (7-3) from above-mentioned Equations (3) and (6).

25

$$\begin{aligned} y(k) &= G0 \cdot u(k-1) \\ &= \sum_{i=1}^n u(k-i) \cdot \text{imp1}(i) + \sum_{i=1}^n u(k-i)^3 \cdot \text{imp3}(i) + \sum_{i=1}^n u(k-i)^5 \cdot \text{imp5}(i) \quad (7-1) \\ &= u(k-1) \cdot \text{imp1}(1) + u(k-1)^3 \cdot \text{imp3}(1) + u(k-1)^5 \cdot \text{imp5}(1) \end{aligned}$$

$$+ \sum_{i=2}^n u(k-i) \cdot \text{imp1}(i) + \sum_{i=2}^n u(k-i)^3 \cdot \text{imp3}(i) + \sum_{i=2}^n u(k-i)^5 \cdot \text{imp5}(i) \quad (7-2)$$

$$= u(k-1) \cdot \text{imp1}(1) + u(k-1)^3 \cdot \text{imp3}(1) + u(k-1)^5 \cdot \text{imp5}(1) + hf(i) \quad (7-3)$$

where the Equation (7-2) represents terms (k-1) apart from other terms.

- 5 Further, in the Equation (7-3), terms of Σ is replaced with $hf(i)$. Then, when $u(k-2)$, $u(k-3)$, $u(k-4)$, ... are determined, $hf(i)$ in the Equation (7-3) is determined and $u(k-1)$ satisfying the Equation (7-3) is determined. That is, this $u(k-1)$ is a predistortion signal to be outputted when $i = 1$.

The distortion compensation circuit according to the present
10 invention uses lookup tables for obtaining the $hf(i)$. More specifically, lookup tables are prepared correspondingly to time intervals τ shown in Fig. 3, respectively, wherein each lookup table stores a value of $\text{imp}(i) \times u(k-i)$ at each time interval τ . In other words, the distortion compensation circuit according to the present invention stores, in the
15 lookup table, values obtained by previously multiplying, by $\text{imp}(*)$, possible values in the entire voltage range of the input signal voltage $u(t)$ to be inputted to the power amplifier. Then, changing an access point of each lookup table in accordance with the input value provides the value obtained by multiplying the input value by the impulse response. Fig. 5
20 shows an example of data in the lookup table used in the distortion compensation circuit according to the present invention.

Further, the distortion compensation circuit according to the present invention also uses a lookup table for obtaining $u(k-1)$ satisfying the Equation (7-3).

- 25 In addition, the distortion compensation described above is effected with respect to amplitude of the signal. However, the distortion compensation circuit according to the present invention also provides distortion compensation in the phase direction in the same manner. That is, if AM/PM characteristics are different at every frequency, phase
30 compensation is represented by replacing $\text{am}(*)$ in the Equation (5) with

those at obtained from frequency dependency of AM/PM as follows:

$$Phout(fi) = pm1(fi) \cdot Vin(fi) + pm2(fi) \cdot Vin^2(fi) + pm3(fi) \cdot Vin^3(fi) \quad (8)$$

5 where $Phout(fi)$ represents a phase component of the output signal of the power amplifier, and $pm(*)$ is similar to $am(*)$ in the Equation (5) and obtained from the frequency characteristic of AM/PM. Using $pm(*)$ in the Equation (8) provides $Pho(t)$ representing variation in $Phout(fi)$ in a time base. The distortion compensation in the phase direction is
10 provided from $Pho(t)$ by shifting the phase of the input signal to the power amplifier by that $PhPD(t) = -Pho(t)$.

Application Example of Distortion Compensation Circuit

Fig. 6 shows a general structure of a transmission section (transmission apparatus) of a portable telephone terminal to which the
15 above-described distortion compensation circuit according to the present invention is applied.

In Fig. 6, the baseband section 1 supplies an in-phase component signal BI and a quadrature component signal BQ to a quadrature modulation section 2. The in-phase component signal BI is supplied to
20 one mixer 11 in the quadrature modulation section 2. The quadrature component signal BQ is supplied to the other mixer 12 in the quadrature modulation section 2. The mixer 11 is supplied with an oscillation signal from an oscillator 14, and the other mixer 12 is supplied with a signal phase-shifted by 90 degrees ($\pi/2$). Thus, the mixer 11 frequency-converts
25 the in-phase component signal BI with the oscillation signal, and the other mixer 12 frequency-converts the quadrature component signal BQ with the 90-degree phase-shifted oscillation signal. An adder 13 adds these output signals of the mixers 11 and 12 to each other and supplies its output as an output of the quadrature modulation section 2 to a distortion
30 compensation section 3 as its input signal PDin.

The distortion compensation section 3 generates the predistortion

signal PDout for compensating distortion in the direction of the amplitude of the power amplifier at the final stage and the predistortion signal PhPD for compensating distortion in the phase direction by changing access points of the above mentioned lookup table in accordance
5 with the input signal PDin. The predistortion signal PDout in the amplitude direction is supplied to a bandpass filter 4 and the predistortion signal PhPD in the phase direction is supplied to a phase shift section 8.

The bandpass filter 4 removes EMI components included in the
10 predistortion signal PDout for compensation in the amplitude direction. The output signal of the bandpass filter 4 is supplied to a mixer 5. The mixer 5 is further supplied with an oscillation signal from a radio frequency oscillator 6. Thus, the mixer 5 frequency-converts the output signal of the bandpass filter 4 with the oscillation signal from the radio
15 frequency oscillator 6. The radio frequency signal from the mixer 5 is supplied to a bandpass filter 7. The bandpass filter 7 removes EMI components included in the radio frequency signal outputted from the mixer 5. The output signal of the bandpass filter 7 is supplied to the phase shift section 8.

20 The phase shift section 8 is supplied with the predistortion signal PhPD for compensation in the phase direction from the distortion compensation section 3 as a phase shift control signal to phase-shift the output signal from the bandpass filter 7. That is, the phase shift section 8 outputs a predistortion signal PAin for compensating distortion both in
25 amplitude and phase directions in the power amplifier 9. Thus, the output signal PAout of the power amplifier 9 is distortion-compensated. The output PAout of the power amplifier 9 is supplied to a rear stage structure (not shown) through a terminal 10.

Here, since the portable telephone terminal according to the
30 embodiment obtains the predistortion signal PDout for compensation in amplitude direction and the predistortion signal PhPD for compensation

in phase direction by accessing the lookup tables in accordance with the input signal PDin, this embodiment particularly takes the following points into consideration.

The oscillation frequency of the oscillator 14 in the quadrature modulation section 2 in Fig. 6 is relative low such that it is about twice an envelope frequency of signals BI and BQ from the baseband section 1. The reason is as follows:

Generally, frequency conversion operation with a mixer or the like also generates an image component, so that the occupied frequency bandwidth becomes twice that of the original signal (the output of the quadrature modulation section 2 in this embodiment). In this case, the memory effect to be considered occurs at a frequency band two or more times the envelope frequency. Thus, the distortion compensation should be made in consideration of this frequency band. However, because the distortion compensation according to the present embodiment, as described above, requires the digital process such that lookup tables are accessed in accordance with the input signal PDin, it is difficult to process data at such a high frequency. Then, in this embodiment, the oscillation frequency of the oscillator 14 is made low such that it is about twice the envelope frequency of the signals BI and BQ to simplify the digital process.

Further, the distortion compensation according to the present embodiment generates the predistortion signal with impulse response from the signal modulated as mentioned above. During this, values stored as impulse response in the lookup table are derived from impulse responses of $am1'(f)$, $am3'(f)$, and $am5'(f)$ shifted, as shown in Fig. 7, to a low frequency region from $am1(f)$, $am3(f)$, and $am5(f)$ in Fig. 2. Fig. 7 is a drawing illustrating simulation to check the distortion compensation effect according to the present invention. This simulates an example in which the oscillation frequency of the oscillator 14 is set to the frequency (1.3 MHz) twice the envelope frequency ($= 1.2288 \text{ MHz}/2$) of the signals

BI and BQ from the baseband section 1. That is, if the radio frequency is $1.3 \text{ MHz} \times 6 = 7.8 \text{ MHz}$, the memory effect of the power amplifier in Fig. 6 is represented by $am1(f)$, $am2(f)$, and $am5(f)$ near 7.8 MHz. However, the distortion compensation section 3 according to the present
5 embodiment accesses the lookup table with $am1'(f)$, $am3'(f)$, and $am5'(f)$ shifted at a frequency range near 1.3 MHz from $am1(f)$, $am2(f)$, and $am5(f)$ in parallel.

Configuration of Distortion Compensation Section

The distortion compensation section 3 according to the present
10 embodiment will be described about structure for generating predistortion signals by accessing to lookup tables in accordance with the input signal PDin. Fig. 8 illustrates an example of a distortion compensation section 3 according to this embodiment.

The distortion compensation section 3, shown in Fig. 8, includes
15 an A/D (analog to digital) converter 32 for digitizing the input signal PDin from the quadrature modulation section 2, a multiplier 33 for multiplying the output of the A/D converter 32 by a constant G_0 , a subtractor 41 supplied with the output of the multiplier 33, a lookup table (power value data output means) 42 supplied with at least a part of the output data of
20 the subtractor 41 as address data for accessing the lookup table 42, a delay block including $N-1$ delays 45_1, 45_2, ... for successively delaying the output of the lookup table 42 by τ , respectively (N is a natural number and in this embodiment, $N = 3$ or 4), N table blocks 46_1, 46_2, ... supplied with the output data successively delayed by τ (output data of the lookup
25 table 42 or output of the delay 45_1, or 45_2 ...), an adder 44 for summing output data of the these table blocks 46_1, 46_2, ... to supply the result to the subtractor 41, and a D/A (Digital to Analog) converter 43, which are provided as an amplitude compensation section 34 for outputting the predistortion signal PDout from the output of the D/A converter 43
30 through a terminal 36 for compensation in the amplitude direction.

The first stage of the table block 46_1 among the respective table

blocks 46_1, 46_2, ... in the amplitude compensation section 34 is supplied with at least a part of the output data of the lookup table 42 as address data for accessing the tables. The table block 46_1 includes M lookup tables 61_1, 61_2, 61_3, ..., each including memory elements and outputting a piece of data from previously stored data in accordance with the address data (M is a natural number and M = 4 in this embodiment), and an adder 47 for adding the output values of the lookup tables 61_1, 61_2, 61_3, ... to each other. The output data of the adder 47 is supplied to the adder 44. The second stage of table block 41_2 has the same structure as the first stage of table block 46_1, but is supplied as table access address data with the output data of the delay 45_1 which is delayed by τ from the output data of the lookup table 42. The table blocks 46_3, ... (not shown) after the second stage of table block 46_2 are similarly provided, wherein the output data of the lookup table 42 is successively delayed by τ with the delays 45_2, and successively delayed data pieces are supplied to table blocks 46_3, ... as address data pieces for accessing tables, respectively. This structure corresponds to the amplitude impulse response accumulation additional means.

The lookup tables 61_1, 61_2, 61_3, ... of the table blocks 46_1, 46_2, ... in the amplitude compensation section 34 store values obtained by multiplying past input signal voltages by impulse responses derived from the above-mentioned $am1'(f)$, $am3'(f)$, and $am5'(f)$, respectively. For example, the first stage lookup table 61_1 stores values of $u(k-2) \cdot imp'1(2)$, the second stage of lookup table 61_2 stores $u(k-2)^3 \cdot imp'3(2)$, and the third stage of lookup table 61_3 stores $u(k-2)^5 \cdot imp'5(2)$.

The lookup table 42 stores values of $u(k-1)$ in the solution satisfying Equation (9).

$$G0 \cdot u(k-1) = u(k-1) \cdot imp'1(1) + u(k-1)^3 \cdot imp'3(1) + u(k-1)^5 \cdot imp'5(1) \quad (9)$$

where $imp'1(i)$, $imp'3(i)$, and $imp'5(i)$ represent impulse responses

obtained from $am1'(f)$, $am3'(f)$, and $am5'(f)$, respectively. Further, the constant $G0$, used for multiplying in the multiplier 33, represents a linear gain when the characteristic of the power amplifier 9, of which distortion is to be compensated, is linearized. As mentioned above, the distortion compensation section 3 according to this embodiment uses the output of the lookup table 42 as address data for the lookup tables 61_1, 61_2, 61_3, ... in the table blocks 48_1, 48_2, ... This is because the signal actually inputted to the power amplifier 9 is that pre-distorted, and thus it is necessary to consider the output value of the lookup table 42 as the input signal sequence.

The distortion compensation section 3 with the structure mentioned above converts the input signal $PDin$ supplied at the terminal 31 and multiplying it by $G0$. The multiplied data is supplied to the lookup table 42 for accessing. At least part of the output of the lookup table 42 is supplied to the lookup tables 61_1, 61_2, 61_3, ... in the table blocks 48_1, 48_2, ... for accessing to obtain $u(k-2) \cdot imp1(2)$ or the like in the Equation (7-3). Next, outputs of these lookup tables 61_1, 61_2, 61_3, ... are added to each other at the adder 47 to provide $hf(i)$ in the Equation (7-3). Further, the subtractor 41 subtracts the output data of the adder 47 from the output data of the multiplier 33 ($PDin \cdot G0$) to obtain signal sequence for accessing the lookup table 42. The output of the lookup table 42 is D/A-converted to provide the analog predistortion $PDout$ for compensation in the amplitude direction.

The distortion compensation section 3, as shown in Fig. 8, further includes N table blocks 48_1, 48_2, ... supplied with output data of the lookup table 42 successively delayed by τ with the delays 45_1, 45_2, ..., an adder 50 for adding the outputs of these table blocks 48_1, 48_2, ... to each other, and a D/A converter 51 for D/A-converting the output data of the adder 50, which are provided as a phase compensation section 35.

The first stage of the table block 48_1 among the respective table blocks 48_1, 48_2, ... in the phase compensation section 35 is supplied

with at least a part of the output data of the lookup table 42 as address data for accessing the tables. The table block 48_1 includes M lookup tables 71_1, 71_2, 71_3, ..., each including memory elements and outputting a piece of data from previously stored data in accordance with the address data, and an adder 49 for adding the output values of the lookup tables 71_1, 71_2, 71_3, ... to each other. The output data of the adder 49 is supplied to the adder 50. The second stage of table block 48_2 has the same structure as the first stage of table block 48_1, but is supplied, as table access address data, with the output data of the delay 45_1 which is delayed by τ from the output data of the lookup table 42. The table blocks 48_3, ... (not shown) after the second stage of table block 48_2 are similarly provided, wherein the output data of the lookup table 42 is further successively delayed by τ and successively delayed data pieces are supplied to table blocks 48_3, ... as address data pieces for accessing tables, respectively. This structure corresponds to the phase impulse response accumulation additional means according to the present invention.

In the phase compensation section 35, lookup tables 71_1, 71_2, 71_3, ..., in of table blocks 48_1, 48_2, ... store values derived by changing signs of impulse responses of coefficients p_m (*) shown in the Equation (8), respectively. Thus, the outputs of the lookup tables 71_1, 71_2, 71_3, ... are summed by the adder 49, and the result is D/A-converted by the D/A converter 51 as the predistortion PhPD for compensation in the phase direction, which controls the phase shift in the phase shift section 8 to provide phase compensation.

In the distortion compensation section 3 in Fig. 8, the delay time τ of each of delays 45_1, 45_2, ... represents the interval for digitizing in Fig. 3. It is better that the delay time τ is short. However, processing speeds of the used A/D converter 32, and the D/A converters 43 and 51 determine the lower limit.

Simulation of Distortion Compensation

Study of effects in distortion compensation according to the present invention through simulation was made. The result will be described below. The condition of simulation is the same as that described in Fig. 7. In addition, the delay time τ is determined by the
5 condition of 16 times over-sampling, i.e., $1/19.2 \text{ MHz} = 1.2288 \text{ Mcps} \times 16$.

Fig. 9 shows an example of spectrum of the output signal PAout with the characteristics of the power amplifier described in Figs. 1 and 2 obtaining through simulation. The signal is used in uplinking in an N-CDMA (Narrowband-Code Division Multiple Access), and the chip rate is
10 1.2288 Mcps, and data modulation is OQPSK (Offset Quadrature Phase Shift Keying). The output signal PAout of the power amplifier is 21.5 dBm. The spectrum in Fig. 9 includes the signal component and distortion components as shown. The signal component shows deviation of about 2 dB within the band (= dev). This is one of characteristic of the
15 memory effect and caused from the above-described $\text{am1}(f)$. Further, there is a difference of about 4 dB in magnitude at ACPs, representing distortion (= $\pm 885 \text{ kHz}$, detuning points), between the high and low sides of the signal component ($\text{dlt_acpr} = \text{ACP_high} - \text{ACP_low}$). That is, asymmetrical distortion spectrum with respect to the high and low sides
20 of the signal component is generated. Similarly, this is caused from above-described $\text{am3}(f)$ and $\text{am5}(f)$. Thus, if the power amplifier has a strong memory effect, it shows a characteristic that the output spectrum becomes asymmetrical.

Figs. 10 and 11 show variation in spectrum between distortion
25 compensation results according to two methods. Fig. 10 shows the effect of distortion compensation in an example wherein the AM/AM characteristic of $f = f_3$ and a signal for linearizing, i.e., a predistortion signal obtained in consideration only of the characteristic of the curve po3 without consideration of the memory effect, are used. This spectrum is
30 obtained when the output signal power of the power amplifier 9 is $\text{PAout} = 21.5 \text{ dBm}$. Hereinafter, this predistortion is referred to as "1 point PD".

Similarly, Fig. 11 shows the predistortion according to a preferred example of the present invention using the Equation (7-3). The output signal power of the power amplifier 9 is also $PA_{out} = 21.5$ dBm. Further, the number of times of accumulating the past values used during obtaining $hf(i)$ in the Equation (7-3) is set five. Hereinafter, this predistortion is referred to as "5 step PD". Further, the example in Fig. 9 is referred to as "w/o PD" because there is no distortion compensation. The example of "1 point PD" in Fig. 10 is observed as if the distortion compensation is effective. However, both of the deviation within band dev and asymmetric ACP are not reduced. Further, there is no improvement in distortion at a range where the detuning frequency is large. On the other hand, in the preferred example of "5 step PD" according to the present invention shown in Fig. 11, the deviation within the band dev and asymmetric ACP are eliminated and further, distortion, at the range where the detuning frequency is large, is sufficiently improved.

Next, Figs. 12 and 13 show output power dependencies of ACP_low and ACP_high, respectively, for three cases shown in Figs. 9 to 11. In Figs. 12 and 13, at the ranges where the output signal power of the power amplifier is low, there are no large difference in output dependencies of "1 point PD" and "5 step PD". At the range where the output signal power PA_{out} is greater than 20 dBm, the output dependency of "5 step PD" shows improvement in ACP by equal to or more than 5 dB. At the range where the output signal power of the power amplifier 9 is large, there is no effect of distortion compensation because distortion due to clipping occurs there.

Figs. 14 and 15 show deviation within band dev and asymmetry in ACP, respectively. Fig. 14 shows an output dependency of deviation within band dev. In both examples of "1 point PD" and "w/o PD", there are some remaining deviations within band, and thus there is not improvement. On the other hand, in the preferred example of

predistortion "5 step PD" according to the present invention, there is an effect because the deviation within band dev is suppressed within 1 dB up to about 22 dBm. Here, at the range of more than 23 dBm, the deviation within band dev becomes large, but this is caused by clipping. Fig. 15 shows the output power dependency of asymmetry of ACP. The asymmetry of ACP dlt_acpr is defined as $dlt_acpr = ACP_high - ACP_low$. In Fig. 15, the asymmetry of ACP dlt_acpr of the example "1 point PD" is approximately the same as that of the example of "w/o PD", and thus there is no improvement. On the other hand, the asymmetry of ACP dlt_acpr of the preferred example of "5 step PD" according to the present invention is within 0.5 dB, and thus there is an extremely high improvement.

As described above, the distortion compensation circuit according to the present invention once modulates the baseband to obtain low frequency signals. After this, the distortion compensation circuit accumulates past signal sequence over a plurality of sampling points, and results of multiplication of impulse response of the power amplifier is used for distortion compensation, so that distortion in a power amplifier with a high memory effect can be compensated in distortion over the entire bandwidth accurately.

The above-mentioned description is only for examples according to the present invention. Therefore, the present invention is not limited to the above-described embodiment and thus it is to be understood that changes and variations may be made according to circumstances without departing from the spirit or scope of the present invention.